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(19)



Europäisches Patentamt
European Patent Office
Office européen des brevets



(11) Publication number:

0 185 029 B1

(12)

EUROPEAN PATENT SPECIFICATION

(45) Date of publication of patent specification: 27.01.93 (51) Int. Cl.⁵: H04M 3/50

(21) Application number: 85902303.8

(22) Date of filing: 17.04.85

(86) International application number:
PCT/US85/00689

(87) International publication number:
WO 85/05000 (07.11.85 85/24)

(54) TELEPHONE OPERATOR VOICE STORAGE AND RETRIEVAL SYSTEM.

(30) Priority: 18.04.84 US 601711

(43) Date of publication of application:
25.06.86 Bulletin 86/26

(45) Publication of the grant of the patent:
27.01.93 Bulletin 93/04

(84) Designated Contracting States:
FR

(56) References cited:
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US-A- 4 066 847 US-A- 4 255 618
US-A- 4 277 649 US-A- 4 328 396
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PATENT ABSTRACTS OF JAPAN, vol. 5, no.
153 (E-76)[825], 26th September 1981; & JP-
A-56 84 063

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EP 0 185 029 B1

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Description

The present invention relates, in general, to telephone systems and is particularly directed to a call answering scheme through which a prerecorded response message is returned to the calling party, either through operator control, or automatically, while permitting the operator to remain on-line and have the ability to inject his/her voice into the communication link to the calling party without any detectable change in quality of the voice being received by the calling party.

Operator-assisted telephone service facilities, such as directory assistance, PBX and Toll Service, require that the service operator handle a large number of similar calls over the operator's work period, with the number of daily incoming requests for assistance typically varying from 800 to 1,300, depending upon the time of day and year. Previous attempts have been made to reduce the operator's work time by employing automatical response systems which contain prerecorded messages usually stored on electromechanical storage and retrieval devices (e.g. magnetic tape cassette or disk devices).

An operator-assisted telephone service facility is described in Patent Abstracts of Japan, vol. 5, No. 153, (E-76)[825], 26th September 1981; & JP-A-56 84 063 (Nippon Denki K.K.) 09-07-1981. It discloses an arrangement for providing a message to an incoming caller, comprising: first means for storing messages; and second means, operable by the operator for accessing a response message from said first means in dependence upon answer of the caller being serviced, causing the accessed message to be played back to said incoming caller.

Unfortunately, the prior art arrangements do not alleviate the operator's burden of performing what is effectively a monotonous routine. Differences in voice characteristics (e.g. tonal quality, accent, gender) tend to create a negative customer reaction.

An additional problem is the fact that incoming calls are placed by customers whose service requests to them are unique and, as such, the customers expect the service provided by the operator to be helpful, courteous and efficient. If, however, the customer's call is answered by a mechanical-sounding prerecorded response message prepared by one person's voice and then followed by the voice of the operator, which not only is different from that of the intercepting response message but conveys a tone that is less than customer-courteous to the caller, it can readily be appreciated how the caller may be confused and often disappointed in the service.

For overcoming the above drawbacks, there is provided an arrangement according to claim 1 and

a method according to claim 11. They reduce the monotonous routine of answering similar types of calls while avoiding drawbacks of conventional automatic response systems, such as those mentioned above, is satisfied by a telephone operator voice storage and retrieval system that is capable of presenting to the customer (caller) a response message (selected from a library of approved answering phrases that have been determined to be the most effective for the inquiries received) in the actual voice of the operator that is on duty at the time. At the same time the present invention permits the operator to follow-up the played-back message with a conversation with the caller, without the caller detecting a difference in the characteristics of the played-back voice and the "live" operator's voice, so that operator voice response message storage and retrieval system of the invention is effectively transparent to the caller.

For this purpose the present invention may employ a voice analyser/synthesizer coupled between a response message memory and an audio interface to the operator's audio equipment (headset). The memory is preferably a high speed/high density semiconductor RAM (such as one contained in the form of a modular cartridge) into which response messages read aloud by the operator are stored after being digitized by the voice analyzer, during the record mode of the system. After a series of approved response messages, as prepared by the on-duty operator, having been stored, the system is ready for use in responding to incoming calls. In this (playback) mode, incoming calls are answered, either by the operator or automatically and, in accordance with the type of call received, the appropriate operator's voice-enunciated response message that had been previously entered into memory is accessed. Then, via the voice synthesizer and an audio interface, that message (in the operator's own voice) is played back to the caller. When the caller speaks again, the operator, who has been on-line the entire time but has been relieved of the need to actually recite the response message, now proceeds to converse with the caller. The audio interface contains automatic level control circuitry which ensures that there is effectively no difference in the audio level of the recorded voice played back to the caller and the "live" voice spoken by the operator. As a result, because recorded message is in the operator's own voice and both live voice and played-back voice are coupled over the same signal flow path, the storage and retrieval system is effectively listener transparent. This listener transparency is an especially attractive feature of the present invention, as it prevents the operator's "live" voice from confusing the caller, as, for example, would be the case if the response message was given in

a female voice abruptly followed by the voice of a male operator handling the remainder of the live call, or releasing a portion of the call to another voice generated by a conventional automatic response system.

The audio interface portion of the system provides full duplex voice transmission (and level control) capability, allowing the operator to hear the caller's voice regardless of line or equipment variations or caller's idiosyncrasies in telephone usage. Advantageously the system makes use of microelectronics signal processing and storage components making it adaptable with a variety of telephone systems and readily intercoupled with an existing operator's console.

BRIEF DESCRIPTION OF THE DRAWINGS

Figure 1 is a simplified block diagram showing of the telephone operator voice message storage and retrieval scheme of the present invention;

Figure 2 is a schematic block diagram of the digitized voice storage and retrieval unit of the scheme shown in Figure 1; and

Figure 3 is a schematic diagram of the audio interface unit of the scheme shown in Figure 1.

DETAILED DESCRIPTION

Before describing, in detail, the particular improved telephone operator voice storage and retrieval system in accordance with the present invention, it should be observed that the invention resides primarily in a novel structural combination of conventional data/signal processing components and communication circuits, and not in the particular detailed configurations thereof. Accordingly, the structure, control and arrangement of these conventional components and circuits have, for the most part, been illustrated in the Drawings by readily understandable block representations and schematic diagrams, which show only those specific details that are pertinent to the present invention, in order not to obscure the disclosure with structural details which will be readily apparent to those skilled in the art having the benefit of the description herein. Thus, for example, multi conductor busses are simplified and power supply terminals and leads have been omitted for clarity. In addition, various portions of an electronic data processing system have been appropriately consolidated and simplified in order to emphasize those portions that are most pertinent to the invention. Thus, the block diagram illustrations of the Figures do not necessarily represent the mechanical structural arrangement of the exemplary system, but are primarily intended to illustrate the major structural

components of the system in a convenient functional grouping, whereby the present invention may be more readily understood.

Referring now to Figure 1 of the drawings, there is shown a generalized block diagram of the telephone operator voice storage and retrieval system of the invention which is to be associated with an operator-assisted telephone service facility. Such a facility is coupled to a link 50 which includes both voice and call identification lines for handling and responding to requests from a caller or customer 70. The tip and sleeve portions of the telephone link are coupled to an audio interface unit 30 (to be described in detail below with reference to Figure 3), while those links indicating the type of incoming call are coupled to a detector 40. It is to be observed here that both the voice and type of call information have been shown as a single link 50 of Figure 1 for purposes of simplifying the illustration and description. The actual signal conveying conductors and detection circuitry for establishing the type of call are conventional and need not be described here for an understanding of the invention.

Call-type detector 40 may be of conventional configuration employing a bank of indicators monitored by an operator 10 who, via a switch panel interface, selects an appropriate code for identifying the type of response message to be returned to the caller 70. This would normally involve the operator monitoring an optical read-out panel of call-type detector 40 and then, via a switch panel interface, causing the playback of a stored response message, such as from a magnetic tape cassette. Rather than have the operator perform this task, however, it is possible to employ a bank of associated detectors, such as opto-electronic detectors, coupled with the indicator unit of the call-type detector 40 of the telephone service facility of interest, which supplies a set of codes over a link 42 to a message storage and retrieval system 20, to be described below with reference to Figure 2. In other words, the type of call being detected may be monitored manually by the operator and the information identifying the type of call coupled to the message storage and retrieval system 20 by an operator switch panel interface, or it may be handled automatically without operator intervention.

The operator 10 is also coupled to the audio interface unit 30 by way of his/her headset, so that an on-line, "live" voice communication may be established between the operator 10 and the caller 70. Pursuant to the present invention, however, the initial conversation/response voice interface between the telephone service facility and the caller 70 is provided from the message storage and retrieval unit 20 without the need for the operator 10

to speak directly to the caller 70. As mentioned previously, this not only provides the intended relief for the operator 10, but ensures that the caller is supplied with the correct response message voice reply (i.e. a preestablished customer-oriented and optimized response that has been previously recorded by the operator in his/her own voice so that the message is pleasing and informative to the listener).

Audio interface unit 30, in addition to providing on-line communication capability between the operator 10 and the caller 70, serves to provide automatic voice level adjustment for all audio that is coupled to the caller 70, whether that audio be generated from the operator's headset or from the message storage and retrieval unit 20. Audio signal coupling between the audio interface unit 30 and the message storage and retrieval unit 20 is effected over input/output lines 11/21.

Referring now to Figure 2, a schematic block diagram of the message storage and retrieval unit 20 is illustrated as being comprised of a voice input/output unit 12, a call type unit 13 and a control unit 14. Voice input/output unit 12 is coupled to audio interface unit 30 by way of audio input link 11 and audio output link 21. Audio input signals input over link 11 represent voice signals generated by the operator 10 during the response message "record" mode of operation of the system for storing response messages to be later played back to reply to incoming calls. These audio signals are digitized and analyzed within the voice input/output unit 12 and stored in digital form in memory in the control unit 14. Similarly, such stored digital encoded voice messages are read-out from memory in the control unit 14, synchronized and coupled over link 21 to the audio interface unit 30 for playback to a caller 70.

Call type unit 13 effectively comprises a buffer for storing an indication of the type of call for which a reply message is to be generated. Call type unit 13 is coupled to the detector 40, either by way of a switch panel that is operator-controlled or through an automatic detector, such as an opto-electronic detector, which monitors the operator/attendant's telephone facility terminal. The data that is stored in buffer 41 is accessed by a control unit 14 for selecting the appropriate response message that has been stored in memory to be read-out and generated as a reply message to be delivered to the caller 70.

The control unit 14 contains processor, memory and communication bus components for controlling the operation of message storage and retrieval unit 20 as operator-generated response messages are stored and later accessed and delivered to a caller, depending upon the type of call buffered by call type unit 13.

DIGITAL STORAGE AND RETRIEVAL UNIT (Figure 2)

Looking now, in greater detail, at the components of the response message storage and retrieval unit 20 of Figure 2, within the voice input/output unit 12, audio input link 11, which receives analog input signals corresponding to the audio voice supplied by the audio interface unit 30, is coupled to an analog-to-digital converter (ADC) 24. ADC 24 samples and digitizes the audio input signal and couples the quantized-encoded sample values of a shift register 31. Shift register 31 serializes out the digitized voice signal samples supplied by ADC 24 over link 33 to a voice analyzer/synthesizer 32. (As mentioned previously, each of the components employed in the present invention is well known, so that no detailed description thereof need be supplied here. For example, the voice analyzer/synthesizer unit 32 may be a commercially available unit from OKI Semiconductor).

For providing the sampling and communication timing among units 24, 31 and 32, a clock source is coupled over link 23 to each of ADC 24, shift register 31 and analyzer/synthesizer 32. It is also coupled to a divider 25 to reduce the frequency for accessing and controlling a load or shift select circuit 26 which is coupled to ADC 24 over link 27 and to shift register 31 over link 28. Shift select logic unit 26 provides the appropriate digitized transfer coupling between ADC 24 and shift register 31, in a customary fashion. A separate clock is coupled over link 34 to analyzer/synthesizer 32 for providing the control of the read-out analyzer/synthesizer 32.

The analog voice that is supplied from synthesizer portion of unit 32 is coupled over link 21 to the audio interface unit 30 in Figure 3, to be described below. The digitized voice components supplied from the analyzer portion of unit 32 are coupled as digitized voice data over interface 35 to a buffer 36. Similarly, the synthesizer portion receives digitized voice data from an output buffer 37 to which voice message words read-out of memory stored in the control unit 14 are coupled, as will be described below.

Each of buffers 36, 37, 41 and a buffer 53 is coupled to data bus 45 within the control unit 14. Buffer 53 is coupled over link 43 to audio-interface unit 30 (Figure 3) and stores a voice detection signal coupled thereto from unit 30. Data bus 45 includes communication highway conductors for conveying both data and a portion of the address signals to be employed within the control unit 14 for accessing the various components thereof. In the exemplary embodiment, data bus 45 may be a commercially available Z-80 type bus structure.

Control unit 14 is an intelligent-based unit under control of microprocessor 61, which is coupled to data bus 45, to an address bus 52 and to a control bus 62 in a conventional manner. The clock source for microprocessor 61 is coupled over link 65.

In addition to microprocessor 61, control unit 14 includes a response message memory 51, such as a static RAM in the form of a pluggable modular cartridge, and a program store unit 46. Each of memories 51 and 46 is coupled to data bus 45 and to address bus 52 for access by microprocessor 61. Program store unit 46 stores the instruction set for controlling the operation of the storage and retrieval unit and the manner in which it interfaces with the remainder of the system. As that instruction set may take on a number of forms depending upon the desires of the programmer, it will not be described in detail here. Instead, the operational scenario carried out by the program stored in memory 46 will be described, in order to provide a more efficient description of the invention.

Control unit 14 also includes a memory select logic unit 54 (consisting of combinational logic) through which microprocessor 61 is selectively accesses response message storage RAM 51 or program store memory 46 via links 56 and 55, respectively. In addition, an input/output select unit logic unit 44 is employed for selectively accessing the contents of buffers 36, 37 and 41 during the operation of the system. Each of select logic units 54 and 44 is accessed by control signals on control bus 62 and address bus 52 from microprocessor 61.

As mentioned previously, call type unit 13, which comprises buffer 41, may be coupled to the control terminal of a telephone service facility either through the operator's control switch panel for manually controlling the response message to be generated to the caller, or through an interface unit which automatically generates control codes to be stored in buffer 41 for accessing the response messages stored in memory 51. In either case, these codes are coupled over a link 42 and stored in buffer 41. As mentioned above, an additional buffer 53 is coupled over link 43. Buffer 53 is monitored by processor 61 for initiating the operation of the storage and retrieval unit of Figure 2. As will be described in detail below, link 43 is coupled to a voice detector within the audio interface unit 30 (Figure 3) which prevents control unit 14 from allocating memory for incoming voice signals until there is actually a voice signal being coupled over link 11. This is employed during the record mode operation wherein the operator will record response messages in her/his own voice for storage in memory 51. Line 43 is monitored by the processor and it is not until the operator actually begins to speak

during this record mode that the processor allocates memory for storing the digitized voice, as opposed to simply beginning successively accessing memory addresses for storing the contents of buffer 36 to which digitized voice data is coupled over link 35 from analyzer 32, without regard to whether ADC 24 is actually receiving voice signals or is waiting for those voice signals from the operator and would thereby be supplying useless data to memory.

Operation

Looking now at the operation of the message storage and retrieval unit, there are two modes of operation of the unit - RECORD and PLAYBACK

RECORD MODE

The record mode is employed by the operator at the telephone service facility to record response messages that will be used during the operator's work period. This task may be advantageously carried out at a redundant storage and retrieval system located in a room or area separated from the operator's work station. As mentioned previously, the memory 51, in which response messages are stored, may be contained in the form of a removable cartridge module as part of the hardware of control unit 14, whereby the operator is able to employ a portable and compact mechanism for generating a library of response messages.

Now, in the course of preparing a file of response messages, predetermined (e.g. supervisor approved/edited) response phrases are recorded by the operator as the operator supplies control signals, as through a switch panel coupled to detector 40, for generating response message designation codes to be stored by buffer 41. The messages themselves are coupled via the audio interface unit 30 over audio input link 11, as described above. The particular response message codes and associated response messages are generated and recorded by the operator in order that the processor 61 will know where to store the response messages in RAM 51. Simply put, each response message will be identified by a binary code coupled from the operator switch panel over link 42 and stored in buffer 41. For example, if the operator were to record a response phrase indicating the name of the facility answering the incoming call and a message of inquiry assistance to be read back to the caller, the corresponding switch over the operator's panel would couple an associated code over link 42 to buffer 41. The operator would then proceed to record the message to be stored in memory. As the operator begins recording the message, the voice detector unit within the audio

interface unit 30 (Figure 3) couples a signal over line 43 to buffer 53. Within the control unit 14, processor 61 cycles through the monitoring of buffers 36, 37, 41 and 53 via data bus 45, address bus 52 and control bus 62, in a conventional fashion. When a first voice bit in buffer 53 is detected, processor 61 begins reading out the contents of buffer 36, which receives the digitized voice from analyzer unit 32, as the voice is digitized by ADC 24, coupled to shift register 31 and serialized over link 33 to voice analyzer 32. These digital codes, representative of the voice message to be stored, are coupled over bus 45 and stored in sequential addresses in RAM 51. When the message is complete, as indicated by lack of a voice detect signal on link 43, the processor terminates the generation of address signals for the storage of the message in RAM 51.

Each additional message is handled in the same way until the operator has completed his/her recording of all the response phrases that are to be used during the operator's work period. The response phrase file contained in RAM 51 may then be removed from the unit employed for message recording for subsequent use in the storage and retrieval system at the operator's work station. Again, because the operator is able to record his/her voice in a time frame approximate that during which the operator will be on-line with incoming calls, characteristics of the voice as stored in memory will be substantially identical to that of the operator when the operator is on-line.

Playback Mode

During the playback mode of operation, when a call type is detected, either through an operator-controlled interface set of switches of the operator's control panel, or automatically, as mentioned above, an access code (corresponding to that originally entered by the operator during the recording of the messages) is coupled over link 42 to buffer 41. Then processor 61 reads the access code stored in buffer 41 and generates the appropriate address signals for accessing the corresponding response message that had been previously stored in memory 51 during the RECORD mode. The response message is coupled from memory 51 over bus 45 and latched in output latch 37. During the read-out cycle provided by input output select logic unit 44, the contents of output buffer 37 are coupled over link 35 to synthesizer portion of voice analyzer/synthesizer 32, so that analog audio signals are supplied over line 21 corresponding to the message of interest as it is accessed from memory 51. These analog audio signals are coupled over link 21 to audio interface unit 30 in Figure 3.

AUDIO INTERFACE UNIT (FIGURE 3)

Referring now to Figure 3, there is shown a schematic block diagram of the audio interface unit 30. As explained previously, this unit is coupled to the telephone voice lines for both the operator's position and the caller 70, and to the storage and retrieval unit 20 in Figure 2. At the operator's position, a dual microphone input is provided from the operator's headset at dual input jack 88. One output of dual input jack 88 is coupled over link 71 to an amplifier 101. The other side of the jack is coupled over link 81 to the sleeve lead of the telephone line at the operator position. The other half of dual input jack 88 has one line 83 grounded while the other line 82 is coupled to the sleeve portion of the telephone link to the caller 70. Amplifier 101 amplifies the operator's input voice signal from the operator's microphone to the appropriate level for the circuitry of the audio interface unit and couples that signal over line 102 to an automatic level control amplifier 103. Line 102 is also coupled to the output of a low pass filter 96 the input of which is coupled to a switch 91 through link 95. Switch 91 is coupled to the audio output link 21 from the storage and retrieval unit 20 in Figure 2. As mentioned above, link 21 supplies the read-out or synthesized voice that had been previously stored in memory. During the PLAYBACK mode, this voice signal is coupled over link 21 to terminal 93 and over switch link 94 to low pass filter 96. Thus, there is an effective summation of the operator's voice and the audio output over link 102 to the input of automatic level control amplifier 103. During the RECORD mode, there is no output over link 21 to the audio interface unit, so that switch arm 94 is coupled to terminal 92, which is grounded.

The output of automatic level control amplifier 103 is coupled via link 105 to a current-voltage converter 104, terminal 106 of switch 115 and amplifier 115. Current-voltage converter may comprise a transistor amplifier, the collector of which is coupled over link 73 to one input of a diode bridge circuit 74. The other input is coupled to ground. Output 75 of bridge circuit 74 is coupled to the tip lead of the telephone link to the operator position 10 while output 76 is coupled to the tip lead of the telephone link to the caller 70.

Thus, via the dual microphone input jack 88, audio input link 21 and outputs 75 and 76 of bridge circuit 74, both the caller and the operator's headset are coupled in parallel to receive the voice from both the operator's microphone input and the voice from the audio output from the synthesizer.

As mentioned above, output link 105 from automatic level control amplifier 103 is coupled to terminal 106 of switch 115. Like switch 91, switch 115

has PLAYBACK and RECORD mode positions. During the PLAYBACK mode, switch arm 111 of switch 115 is coupled to terminal 107 (floating), so that there is no output coupled over link 112 from terminal 108 to a low pass filter 113. The output of filter 113 is coupled via amplifier 114 to audio input link 111 to the voice analyzer portion of the storage and retrieval unit 20 shown in Figure 2. In the RECORD mode, however, during which the operator is reading aloud a response message to be stored in memory, switch arm 111 is coupled to terminal 106, so that the operator's voice is coupled over link 11 to the storage and retrieval unit.

A further component of the audio interface unit is a first voice detector which comprises an amplifier 116 coupled to link 105 and a monostable multi-vibrator (or one-shot) 117 coupled to the output of amplifier 116. The output of one-shot 117 is coupled to link 43 to provide a first voice indication. Simply put, amplifier 116 and one-shot 117 monitor line 105 for a voice signal and then supply a trigger signal over line 43 to be stored in buffer 41, as mentioned previously.

Operation

As mentioned above, the audio interface unit operates in either RECORD mode or a PLAYBACK mode.

RECORD MODE

During the RECORD mode, the operator is reading a message to be analyzed and converted into digital signals for storage in memory in the storage and retrieval unit 20 (Figure 2). During this mode, each of switches 91 and 115 is switched to the RECORD mode position by the operator. The operator then speaks the messages into the microphone of his/her headset. The analog audio signals are coupled over link 71, amplified by amplifier 101 and then level-controlled by amplifier 103. The output of amplifier 103 is coupled via switch 115 to low pass filter 113 and finally to amplifier 114 for application to the audio input link 11 which is coupled to the digital storage and retrieval portion of the system. The operator is able to listen to his/her voice by coupling of tip leads 75 and 76 and sleeve leads 81 and 82 to the operator's position, as shown in Figure 3.

PLAYBACK MODE

When the operator has finished recording all of the response messages that will be employed during the operator's tour of duty, the operator changes the positions of switches 91 and 115 to the PLAYBACK position (the position shown in Fig-

ure 3). In this position, switch 91 couples any audio output from the storage and retrieval unit to automatic level control amplifier 103 for application over link 73 to both output tip lead 75 and output tip lead 76. The voice signal is not coupled to the audio input link 11, since switch is effectively open. Any voice signal spoken by the operator into the operator's microphone in his/her headset is coupled via amplifier 101 and link 102 to amplifier 103 and fed downstream to both the operator and caller positions in exactly the same manner as the audio output from the voice synthesizer in the storage and retrieval unit as coupled over link 21. Thus, the operator is able to listen to the voice message that is being played back from the storage and retrieval portion of the system and both his/her own voice when he/she speaks during further conversation with the caller. This monitoring capability and the fact that the operator is listening to his/her own previously recorded voice offers a significant improvement over conventional automatic response systems.

More specifically, the voice that is heard by the caller 70, whether it be the operator's own voice supplied from the operator's microphone over input link 71, or from the audio output from the synthesizer over input line 21, appears to be the same voice, in terms of quality and amplitude. The quality is the same because the voice message is a message in the voice of the operator who is actually providing the service at the telephone facility handling the caller's incoming call. In addition, because both the operator's voice signal supplied from his/her microphone and the synthesized voice signal supplied from the digital storage equipment are coupled to the same amplification and level adjustment circuitry, there is no sharp inflection or level change between the two voice signals. Thus, the storage and retrieval and audio processing circuitry is effectively listener transparent.

An additional advantage of the present invention is the fact that when a response message is initially played back to the caller, the operator, while resting, is listening to that voice response message. When the operator had previously read that message from a carefully prepared and supervised text during the record mode of the system, the operator did so in a clear and courteous manner and pleasant tone while fresh and interested. Because the operator now hears his/her tone of voice being played back in such a manner, the operator is being effectively psychologically stimulated to follow his/her own voice and not create a discontinuity to the listener. Thus, not only does the present invention prevent the above-mentioned confusion problem from arising, but it effectively provides guidance for the operator in the manner in which the operator should speak to the caller, thus

acting as a voice refresher.

Claims

1. For use with an operator-assisted telephone service facility in which on-line participation by an on-line operator is required for effectively enabling an incoming caller to reach a called destination, an arrangement for providing a response message to said incoming caller accessing said facility, comprising:
 first operational means (14) or storing at least one prescribed response message, to be played back, and second means (13) operable for accessing a response message from said first operational means causing said accessed response message to be played back to said incoming caller, characterized in that said first operational means are arranged for said prescribed response message to effectively correspond to the voice of the operator who is on-line with and services incoming calls; and in that said second operational means are coupled to said first operational means and operable in conjunction with the on-line operator's servicing of an incoming call for accessing said response message in dependence upon information contained within said incoming call being serviced by said on-line operator and which is representative of the type of call to which said incoming call corresponds.
2. An arrangement according to claim 1, wherein said second operational means includes means for providing a path for coupling voice messages from said operator to said incoming caller in addition to coupling said played-back message from said first operational means to said incoming caller.
3. An arrangement according to claim 1, wherein said second operational means includes means for enabling the simultaneous coupling to said incoming caller of a voice message from said operator and of a played-back message from said first operational means.
4. An arrangement according to claim 3, wherein said second operational means includes means for adjusting, in effectively the same manner, at least one prescribed characteristic of each of said voice message from said operator and said played-back message from said first operational means.
5. An arrangement according to claim 4, wherein said adjusting means includes means for adjusting the amplitude of messages coupled by

said second operational means to said incoming caller.

6. An arrangement according to claim 1, wherein said second operational means includes means for coupling a voice message from said operator to be stored by said first means through a common signal-flow path for a response message played back as a voice message to said incoming caller.
7. An arrangement according to any one of claims 1-6, wherein said first operational means includes means for initiating the storage of a response message as voiced by said operator in response to the voicing of said message by said operator.
8. An arrangement according to claim 1, wherein said second operational means includes means for enabling said operator to monitor said incoming call and the response message played back in response thereto while also being able to effect a voice communication with said incoming caller.
9. An arrangement according to claim 8, wherein said second operational means includes means for adjusting, in effectively the same manner, at least one prescribed characteristic of each of a voice communication from said operator and a response message played back to an incoming caller.
10. An arrangement according to any one of claims 1-9, wherein said first operational means stores a plurality of response messages, each of which, when played back, effectively corresponds to the voice of the operator who is on-line with and services incoming calls.
11. For use with an operator-assisted telephone service facility on which on-line participation by an on-line operator is required for effectively enabling an incoming caller to reach a called destination, a method of supplying response messages to an incoming caller accessing said facility comprising the step of:
 - (a) storing at least one response message, to be played back,
 - (b) in conjunction with the servicing of an incoming call, accessing said at least one response message and causing said response message to be played back to said incoming caller,
 characterized in that during step (a) response messages are stored which, when

played back effectively correspond to the voice of the operator to be attending to the servicing of incoming calls; and, during step (b), said response message is accessed in dependence upon information contained within said incoming call being serviced by said on-line operator and which is representative of the type of call to which said incoming call corresponds.

12. A method according to claim 11, characterized in that step (b) includes simultaneously enabling the coupling to said incoming caller of a voice message from said operator and playing back an accessed response message to said incoming caller.
13. A method according to claim 11 or 12, characterized in that it further includes:
(c) enabling the coupling of voice messages for said operator to said incoming caller in addition to causing said response message to be played back to said incoming caller.
14. A method according to claim 13, characterized in that step (c) includes adjusting the amplitude of messages coupled to said incoming caller.
15. A method according to any one of claims 11-14, characterized in that step (a) comprises the steps of
(a-1) converting a response message as voiced by said operator into digitally encoded format, and
(a-2) storing a response message as digitally encoded in step (a-1), and in that step (b) includes reconverting a digitally encoded response message into a voice message played back to said incoming caller.
16. A method according to claim 11, characterized in that step (b) includes coupling a voice message from said operator to be stored through a common signal-flow path for a response message played back as a voice message to said incoming caller.
17. A method according to claim 16, characterized in that step (b) includes simultaneously enabling the coupling to said incoming caller of a voice message from said operator and a response message played back as a voice message.
18. A method according to claim 16, characterized in that step (b) includes adjusting at least one prescribed characteristic of each of said voice message from said operator and said played

back message in effectively the same manner.

19. A method according to claim 11, characterized in that step (a) includes initiating the storage of a response message as voiced by said operator in response to the voicing of said message by said operator.
20. A method according to claim 11, characterized in that step (a) comprises the step of storing a plurality of prescribed response messages which, when played back effectively correspond to the voice of the operator to be attending to the servicing of incoming calls, and in that step (b) includes the step of, in conjunction with the servicing of an incoming call, accessing a response message selected in accordance with the type of incoming call being serviced and causing said selected response message to be played back to said incoming caller.
21. A method according to claim 20, characterized in that step (b) includes enabling said operator to monitor said incoming call and the response message played back in response thereto while also providing the capability of effecting a voice communication with said incoming caller.
22. A method according to claim 2, characterized in that step (b) comprises the step of adjusting, in effectively the same manner, at least one prescribed characteristic of each of a voice communication from said operator and a response message played back to an incoming caller.

Patentansprüche

1. Anordnung zum Verwenden bei einer handvermittelten Fernsprechdienst-Einrichtung, in welcher eine direkt angeschlossene Teilnahme durch einen direkt angeschlossenen Betätiger benötigt wird, um einem Eingangs-Anrufer wirksam zu ermöglichen, ein angerufenes Ziel zu erreichen, zum Vorsehen einer antwortabhängigen Mitteilung an den Eingangs-Anrufer, der Zugriff zur Einrichtung hat, die aufweist:
erste Betriebsmittel (14) zum Speichern wenigstens einer festgelegten, antwortabhängigen Mitteilung, die wiedergegeben wird, und zweite betriebsfähige Mittel (13), um Zugriff auf eine antwortabhängige Mitteilung vom ersten Betriebsmittel zu haben, die veranlassen, daß die in Zugriff genommene, antwortabhängige Mitteilung an den Eingangs-Anrufer wiedergegeben wird,

dadurch gekennzeichnet,

daß die ersten Betriebsmittel für die festgelegte, antwortabhängige Mitteilung vorgesehen sind, um wirklich der Stimme des Betätigers zu entsprechen, der direkt mitangeschlossen ist und die Eingangs-Anrufe verwaltet; und daß die zweiten Betriebsmittel mit den ersten Betriebsmitteln gekoppelt sind und betriebsfähig sind in Verbindung mit dem direkt angeschlossenen Verwalten des Betätigers eines Eingangs-Anrufs für einen Zugriff auf die antwortabhängige Mitteilung in Abhängigkeit von der innerhalb des Eingangs-Anrufs enthaltenen Information, die durch den direkt angeschlossenen Betätiger verwaltet wird und die typisch ist von der Art des Anrufs, welchem der Eingangs-Anruf entspricht.

2. Anordnung nach Anspruch 1, worin das zweite Betriebsmittel Mittel zum Vorsehen eines Pfades zum Koppeln von Sprach-Mitteilungen vom Betätiger mit dem Eingangs-Anrufer neben dem Koppeln der Wiedergabe-Mitteilung von ersten Betriebsmitteln mit dem Eingangs-Anrufer einschließt.
3. Anordnung nach Anspruch 1, worin das zweite Betriebsmittel Mittel zum Ermöglichen des gleichzeitigen Koppelns einer Sprach-Mitteilung vom Betätiger und einer wiedergegebenen Mitteilung von den ersten Betriebsmitteln mit dem Eingangs-Anrufer einschließt.
4. Anordnung nach Anspruch 3, worin das zweite Betriebsmittel Mittel zum Justieren in wirklich derselben Art und Weise von wenigstens einer festgelegten Eigenart von jeder Sprach-Mitteilung vom Betätiger und der wiedergegebenen Mitteilung von den ersten Betriebsmitteln einschließt.
5. Anordnung nach Anspruch 4, worin das Justiermittel Mittel zum Justieren der Ausschlagweite von Mitteilungen einschließt, die durch die zweiten Betriebsmittel mit dem Eingangs-Anrufer gekoppelt sind.
6. Anordnung nach Anspruch 1, worin das zweite Betriebsmittel Mittel zum Koppeln einer Sprach-Mitteilung vom Betätiger einschließt, damit dort die ersten Mittel durch einen üblichen Signallaufpfad für eine antwortabhängige Mitteilung gespeichert werden, die als eine Sprach-Mitteilung an den Eingangs-Anrufer wiedergegeben wird.
7. Anordnung nach irgendeinem der Ansprüche 1

bis 6,

worin das erste Betriebsmittel Mittel zum Einleiten des Speicherns einer als durch den Betätiger in Worte gefaßten antwortabhängigen Mitteilung in Reaktion auf das In-Worte-Fassen der Mitteilung durch den Betätiger einschließt.

8. Anordnung nach Anspruch 1, worin das zweite Betriebsmittel Mittel einschließt, dem Betätiger zu ermöglichen, sich in den Eingangs-Anruf und die in Reaktion darauf wiedergegebene antwortabhängige Mitteilung einzuschalten, während er auch imstande ist, eine Sprach-Übertragung mit dem Eingangs-Anrufer zu bewerkstelligen.
9. Anordnung nach Anspruch 8, worin das zweite Betriebsmittel Mittel zum Justieren in wirklich derselben Art und Weise wenigstens einer festgelegten Eigenart von je einer Sprach-Übertragung vom Betätiger und eine antwortabhängige Mitteilung einschließt, die an einen Eingangs-Anrufer wiedergegeben wird.
10. Anordnung nach irgendeinem der Ansprüche 1 bis 9, worin das erste Betriebsmittel eine große Anzahl von antwort - abhängigen Mitteilungen speichert, von denen jede, wenn sie weitergegeben wird, wirklich der Stimme des Betätigers entspricht, der direkt mitangeschlossen ist und die Eingangs-Anrufe verwaltet.
11. Verfahren zum Anwenden bei einer handvermittelten Fernsprechklient-Einrichtung, in welcher eine direkt angeschlossene Teilnahme durch einen direkt angeschlossenen Betätiger benötigt wird, um einem Eingangs-Anrufer wirksam zu ermöglichen, ein angerufenes Ziel zu erreichen, zum Liefern antwortabhängiger Mitteilungen an einen Eingangs-Anrufer, der Zugriff zur Einrichtung hat, die die Stufe aufweist vom:
 - a) Speichern wenigstens einer antwortabhängigen Mitteilung, die wiedergegeben wird,
 - b) Zugriff haben in Verbindung mit dem Verwalten eines Eingangs-Anrufs zu wenigstens einer antwortabhängigen Mitteilung und Veranlassen, daß die antwortabhängige Mitteilung an den Eingangs-Anrufer wiedergegeben wird,

dadurch gekennzeichnet,

daß während der Stufe (a) antwortabhängige Mitteilungen gespeichert werden, die, wenn sie wirklich wiedergegeben werden, der Stimme des Betätigers entsprechen, der dem Verwal-

- ten von Eingangs-Anrufen Beachtung schenken muß; und während Stufe (b) die antwortabhängige Mitteilung Zugriff findet in Abhängigkeit von der im Eingangs-Anruf enthaltenen Information, wobei sie durch den direkt angeschlossenen Betätiger verwaltet wird, und die typisch für die Anrufart ist, welcher der Eingangs-Anruf entspricht.
12. Verfahren nach Anspruch 11, **dadurch gekennzeichnet**, daß Stufe (b) Ermöglichen des Koppeln einer Sprach-Mitteilung vom Betätiger und Wiedergeben einer Zugriff gefundenen, antwortabhängigen Mitteilung zum Eingangs-Anrufer gleichzeitig einschließt.
13. Verfahren nach Anspruch 11 oder 12, **dadurch gekennzeichnet**, daß es ferner einschließt.
- c) Ermöglichen des Koppeln einer Sprach-Mitteilung für den Betätiger mit dem Eingangs-Anrufer neben dem Veranlassen, daß die antwortabhängige Mitteilung an den Eingangs-Anrufer wiedergegeben wird.
14. Verfahren nach Anspruch 13, **dadurch gekennzeichnet**, daß Stufe (c) das Justieren der Ausschlagweite der mit dem Eingangs-Anrufer gekoppelten Mitteilung einschließt.
15. Verfahren nach irgendeinem der Ansprüche 11 bis 14, **dadurch gekennzeichnet**, daß Stufe (a) die Stufen aufweist von
- (a-1) Umwandeln einer antwortabhängigen Mitteilung, wie sie durch den Betätiger in Worte in ein digital codiertes Format gefaßt wird, und
- (a-2) Speichern einer antwortabhängigen Mitteilung als in Stufe (a-1) digital codiert und
- daß Stufe (b) Umwandeln einer digital codierten antwortabhängigen Mitteilung in eine Sprach-Mitteilung einschließt, die dem Eingangs-Anrufer wiedergegeben wird.
16. Verfahren nach Anspruch 11, **dadurch gekennzeichnet**, daß Stufe (b) Koppeln einer Sprach-Mitteilung vom Betätiger einschließt, die durch einen üblichen Signalfußpfad für eine antwortabhängige Mitteilung gespeichert wird, die als eine Sprach-Mitteilung dem Eingangs-Anrufer wiedergegeben wird.
17. Verfahren nach Anspruch 16, **dadurch gekennzeichnet**, daß Stufe (b) Ermöglichen des Koppeln einer Sprach-Mitteilung vom Betätiger mit dem Eingangs-Anrufer und eine als Sprach-Mitteilung wiedergegebene, antwortabhängige Mitteilung gleichzeitig einschließt.
18. Verfahren nach Anspruch 16, **dadurch gekennzeichnet**, daß Stufe (b) Justieren wenigstens einer festgelegten Eigenart jeder Sprachmitteilung vom Betätiger und die Wiedergabe-Mitteilung in wirklich derselben Art und Weise einschließt.
19. Verfahren nach Anspruch 11, **dadurch gekennzeichnet**, daß Stufe (a) Einleiten des Speicherns der antwortabhängigen Mitteilung als durch den Betätiger in Worte gefaßt in Reaktion auf das In-Worte-Fassen der Mitteilung durch den Betätiger einschließt.
20. Verfahren nach Anspruch 11, **dadurch gekennzeichnet**, daß Stufe (a) die Stufe zum Speichern einer großen Anzahl von festgelegten, antwortabhängigen Mitteilungen aufweist, die, wenn sie wirklich wiedergegeben werden, der Stimme des Betätigers entsprechen, der dem Verwalten von Eingangs-Anrufen Beachtung schenken muß, und daß Stufe (b) in Verbindung mit dem Verwalten eines Eingangs-Anrufs die Stufe des Zugriffhabens mit einer in Übereinstimmung mit der Art des Eingangs-Anrufs ausgewählten, antwortabhängigen Mitteilung einschließt, die verwaltet und veranlaßt wird, daß die ausgewählte, antwortabhängige Mitteilung dem Eingangs-Anrufer wiedergegeben wird.
21. Verfahren nach Anspruch 20, **dadurch gekennzeichnet**, daß Stufe (b) einschließt, dem Betätiger zu ermöglichen, sich in den Eingangs-Anruf und die in Reaktion darauf wiedergegebene antwortabhängige Mitteilung einzuschalten, während sie ferner die Fähigkeit, eine Sprach-Übertragung mit dem Eingangs-Anrufer zu bewerkstelligen, vorsieht.
22. Verfahren nach Anspruch 2, **dadurch gekennzeichnet**, daß Stufe (b) die Stufe des Justierens in wirklich derselben Art und Weise wenigstens einer festgelegten Eigenart von je einer Sprach-Übertragung vom Betätiger und einer antwortabhängigen Mitteilung aufweist, die dem

Eingangs-Anrufer wiedergegeben wird.

Revendications

1. Agencement, destiné à être utilisé avec un service téléphonique assisté par opérateur dans lequel une participation en ligne d'un opérateur en ligne est nécessaire pour permettre effectivement à une personne qui appelle d'atteindre une destination appelée, pour adresser un message de réponse à ladite personne qui appelle et qui a obtenu l'accès audit service, comprenant:
 - des premiers moyens opérationnels (14) pour stocker au moins un message de réponse prescrit, devant être lu, et des seconds moyens (13) aptes à obtenir un accès à un message de réponse provenant desdits premiers moyens opérationnels et provoquant la lecture dudit message de réponse auquel un accès à été obtenu à ladite personne qui appelle,
 - caractérisé en ce que lesdits premiers moyens opérationnels sont agencés de manière que ledit message de réponse prescrit corresponde effectivement à la voix de l'opérateur qui est en ligne avec et est en charge des appels qui arrivent; et en ce que lesdits seconds moyens opérationnels sont couplés auxdits premiers moyens opérationnels et aptes, en conjonction avec la prise en charge par l'opérateur en ligne d'un appel arrivant, à donner accès audit message de réponse en fonction de l'information contenue dans ledit appel arrivant qui est pris en charge par ledit opérateur en ligne et qui est représentatif du type d'appel auquel correspond ledit appel arrivant.
2. Agencement selon la revendication 1, dans lequel lesdits seconds moyens opérationnels comprennent des moyens pour déterminer un parcours pour le couplage de messages vocaux dudit opérateur à ladite personne qui appelle en plus du couplage dudit message lu à partir desdits moyens opérationnels à la personne qui appelle.
3. Agencement selon la revendication 1, dans lequel lesdits seconds moyens opérationnels comprennent des moyens pour déterminer un couplage simultané avec ladite personne qui appelle d'un message vocal provenant dudit opérateur et d'un message lu provenant desdits premiers moyens opérationnels.
4. Agencement selon la revendication 3, dans lequel lesdits seconds moyens opérationnels comprennent des moyens pour ajuster, effective-

vement de la même manière, au moins une caractéristique prescrite à la fois dudit message vocal provenant dudit opérateur et dudit message lu provenant desdits premiers moyens opérationnels.

5. Agencement selon la revendication 4, dans lequel lesdits moyens d'ajustement comprennent des moyens pour ajuster l'amplitude des messages couplés par lesdits seconds moyens opérationnels avec ladite personne qui appelle.
6. Agencement selon la revendication 1, dans lequel lesdits seconds moyens opérationnels comprennent des moyens pour coupler un message vocal provenant dudit opérateur qui doit être stocké dans lesdits premiers moyens par l'intermédiaire d'un parcours commun de passage de signaux pour un message de réponse lu sous forme d'un message vocal à ladite personne qui appelle.
7. Agencement selon l'une quelconque des revendications 1 à 6, dans lequel lesdits premiers moyens opérationnels comprennent des moyens pour initier le stockage d'un message de réponse exprimé par ledit opérateur en réponse à l'énonciation dudit message par ledit opérateur.
8. Agencement selon la revendication 1, dans lequel lesdits seconds moyens opérationnels comprennent des moyens pour habiliter ledit opérateur à surveiller ledit appel arrivant et le message de réponse lu en réponse à cet appel tout en étant également capable d'établir une communication vocale avec ladite personne qui appelle.
9. Agencement selon la revendication 8, dans lequel lesdits seconds moyens opérationnels comprennent des moyens pour ajuster, effectivement de la même manière, au moins une caractéristique prescrite à la fois de la communication vocale provenant dudit opérateur et d'un message de réponse lu à une personne qui appelle.
10. Agencement selon l'une quelconque des revendications 1 à 9, dans lequel lesdits premiers moyens opérationnels stockent plusieurs messages de réponse, dont chacun, lorsqu'il est lu, correspond effectivement à la voix de l'opérateur qui est en ligne et qui est en charge des appels arrivants.
11. Procédé destiné à être utilisé avec un service téléphonique assisté par opérateur dans lequel

une participation en ligne d'un opérateur en ligne est nécessaire pour permettre effectivement à une personne qui appelle d'atteindre une destination appelée pour fournir des messages de réponse à une personne qui appelle et qui a obtenu l'accès audit service, comprenant les étapes consistant à:

(a) stocker au moins un message de réponse qui doit être lu;

(b) établir un accès en conjonction avec la prise en charge d'un appel arrivant à au moins un message de réponse et provoquer la lecture dudit message de réponse à ladite personne qui appelle,

caractérisé en ce que pendant l'étape (a) sont stockés des messages qui, lorsqu'ils sont lus, correspondent effectivement à la voix de l'opérateur qui est en charge des appels arrivants; et, pendant l'étape (b), est établi un accès avec ledit message de réponse en fonction de l'information contenue dans ledit appel arrivant qui est pris en charge par ledit opérateur en ligne et qui est représentatif du type d'appel auquel correspond ledit appel arrivant.

12. Procédé selon la revendication 11, caractérisé en ce que l'étape (b) comprend l'habilitation simultanée du couplage avec ladite personne qui appelle d'un message vocal provenant dudit opérateur et de la lecture d'un message de réponse auquel a été établi un accès à ladite personne qui appelle.

13. Procédé selon la revendication 11 ou 12, caractérisé en ce qu'il comprend en outre:

(c) l'habilitation du couplage de messages vocaux par ledit opérateur avec ladite personne qui appelle en plus de la détermination de la lecture du message de réponse à ladite personne qui appelle.

14. Procédé selon la revendication 13, caractérisé en ce que l'étape (c) comprend l'ajustement de l'amplitude des messages qui sont couplés avec ladite personne qui appelle.

15. Procédé selon l'une quelconque des revendications 11 à 14, caractérisé en ce que l'étape (a) comprend les étapes consistant à:

(a-1) convertir un message de réponse exprimé par ledit opérateur en un format codé sous forme numérique et

(a-2) stocker un message de réponse codé sous forme numérique à l'étape (a-1), et en ce que

l'étape (b) comprend la reconversion d'un message de réponse codé sous forme numérique en un message vocal lu à ladite personne

qui appelle.

16. Procédé selon la revendication 11, caractérisé en ce que l'étape (b) comprend le couplage d'un message vocal provenant dudit opérateur et qui doit être stocké par l'intermédiaire d'un parcours de passage de signaux commun et destiné à un message de réponse lu sous forme d'un message vocal à ladite personne qui appelle.

17. Procédé selon la revendication 16, caractérisé en ce que l'étape (b) comprend l'habilitation simultanée du couplage avec ladite personne qui appelle d'un message vocal provenant dudit opérateur et d'un message de réponse lu sous forme d'un message vocal.

18. Procédé selon la revendication 16, caractérisé en ce que l'étape (b) comprend l'ajustement d'au moins une caractéristique prescrite à la fois dudit message vocal provenant dudit opérateur et dudit message lu, effectivement de la même manière.

19. Procédé selon la revendication 11, caractérisé en ce que l'étape (a) comprend l'initiation du stockage d'un message de réponse exprimé par ledit opérateur en réponse à l'énonciation dudit message par ledit opérateur.

20. Procédé selon la revendication 11, caractérisé en ce que l'étape (a) comprend l'étape de stockage de plusieurs messages de réponse prescrits qui, lorsqu'ils sont lus, correspondent effectivement à la voix de l'opérateur qui est en charge du service des appels, et en ce que l'étape (b) comprend l'étape consistant, en conjonction avec la prise en charge d'un appel qui arrive, à établir un accès à un message de réponse choisi en accord avec le type d'appel qui est pris en charge et à provoquer la lecture dudit message de réponse sélectionné à ladite personne qui appelle.

21. Procédé selon la revendication 20, caractérisé en ce que l'étape (b) comprend l'habilitation dudit opérateur à surveiller ledit appel et le message de réponse qui est lu en réponse à cet appel tout en fournissant également la possibilité d'établir une communication vocale avec ladite personne qui appelle.

22. Procédé selon la revendication 2, caractérisé en ce que l'étape (b) comprend l'étape d'ajustement, effectivement de la même manière, d'au moins une caractéristique prescrite à la fois d'une communication vocale provenant du-

dit opérateur et d'un message de réponse lu à
une personne qui appelle.

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FIG. 1.

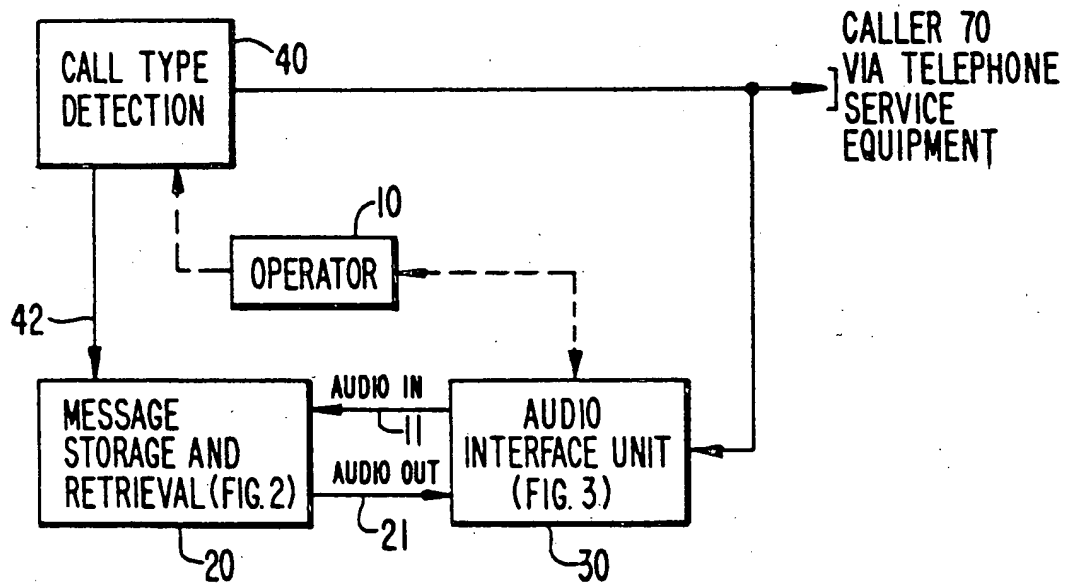


FIG. 2.

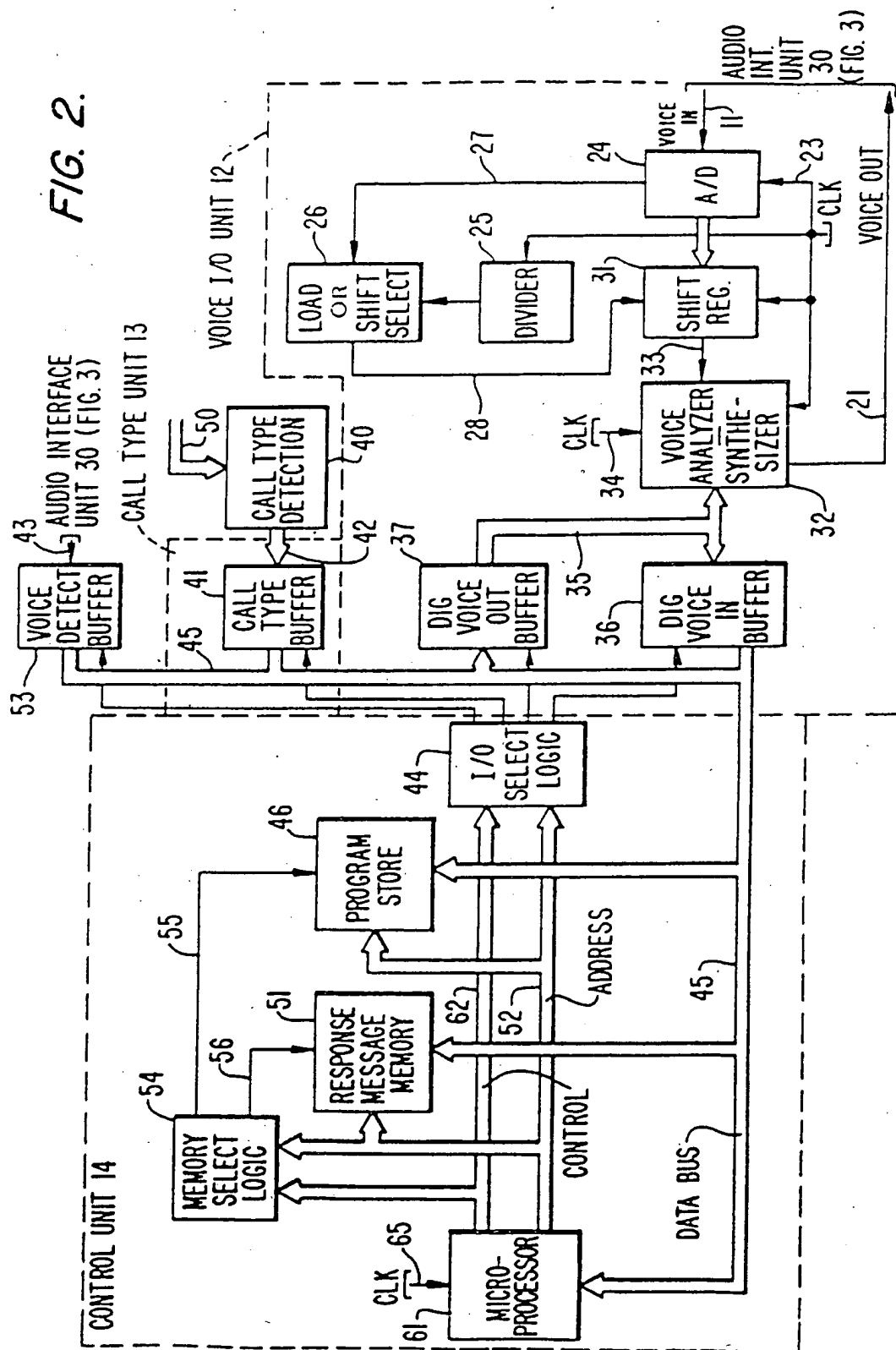


FIG. 3.

